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# EFFICIENT ACCESSING OF A MULTIAACCESS CHANNEL

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## Abstract

A discrete time multiaccess channel is considered where the outcome of a transmission is either "idle", "success", or "collision", depending on the number of users transmitting simultaneously. Messages involved in a "collision" must be retransmitted. An efficient access allocation policy is developed for the case where infinitely many sources generate traffic in a Poisson manner and can all observe the outcomes of the previous transmissions. Its rate of success is .488. Modifications are presented for the cases where the transmission times depend on the transmission outcomes and where observations are noisy.

## I. Introduction

We consider the following model of a multiple access channel. A large number of sources generate messages in a Poisson manner, at a total rate of  $\lambda$  messages per unit of time, starting at time 0. Once a message has been generated, its source can transmit it on a common channel. Transmissions can only start at integer multiples of the unit of time and last one unit of time. If the transmissions from two or more sources overlap, a collision is said to occur, all messages are lost and must be retransmitted at a later time. If only one source transmits, the transmission is successful.

All sources can observe the channel and learn instantaneously whether it is idle, or if a success or a collision has occurred. This common feedback is the only information the sources share. The problem is to find an effective way of using the feedback to schedule the transmission of the messages.

The previous model is an idealization of practical communication systems [1], [2], [3] that have been the object of numerous papers in the communication theory literature [4], [5]. Similar problems have also been treated in control theory journals [6], [7], [8], indeed they are nice examples of distributed control.

The next section will present the basic algorithm and some of its properties. This will be followed by an analysis and optimization. We then show how to modify and analyze the algorithm if the transmission times depend on the transmission outcomes or if the feedback is noisy.

## II. The Basic Algorithm

The algorithm defined below allows the transmissions of the messages on the basis of their generation times, rather than on the basis of the identities of their sources. It has the advantage of being effective no matter the number of sources, even infinite, and is a generalization of the procedure presented in [9].

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The set of messages that are transmitted during the  $n$ th time unit interval are those generated in the time interval  $[y_n, y_n + F(s_n, t_n))$ .  $F$  is a given function (to be optimized below) mapping  $R^+ \cup \{\infty\} \times R^+ \cup \{\infty\}$  into  $R^+$ , with the property that  $F(s, t) < t$ .  $y_0, s_0$  and  $t_0$  are initially equal respectively to 0,  $\infty$  and  $\infty$ , and the  $y_n$ 's,  $s_n$ 's and  $t_n$ 's ( $s_n \leq t_n$ ) are updated by the following rule, where  $F_n = F(s_n, t_n)$ . If a transmission results in

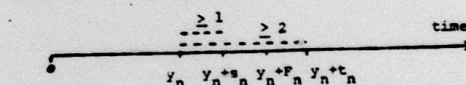
$$\begin{aligned} \text{idle:} & \begin{cases} y_{n+1} = y_n + F_n \\ s_{n+1} = s_n - F_n \\ t_{n+1} = t_n - F_n \end{cases} \\ \text{success:} & \begin{cases} y_{n+1} = y_n + F_n \\ s_{n+1} = t_n - F_n \\ t_{n+1} = \infty \end{cases} \\ \text{collision:} & \begin{cases} y_{n+1} = y_n \\ s_{n+1} = \min(s_n, F_n) \\ t_{n+1} = F_n \end{cases} \end{aligned}$$

Given the outcomes of the  $n$  past transmissions the following can be said about the message generation times:

1. all messages generated in  $[0, y_n)$  have been successfully transmitted
2. the message generation times in  $[y_n, t_n)$  are distributed according to a Poisson process with rate  $\lambda$ , and are independent of the generation times in  $[0, y_n + t_n)$
3. the message generation times in  $[y_n, y_n + t_n)$  are independent of the message generation times in  $[0, y_n)$  and  $[y_n + t_n, \infty)$ , and are distributed according to a Poisson process with rate  $\lambda$ , conditioned on the facts that there are at least one generation time in  $[y_n, y_n + s_n)$  and two generation times in  $[y_n, y_n + t_n)$ .

To prove those assertions, note that they are true initially ( $n=0$ ) and remain true as  $n$  increases. Checking them all is simple, but a rigorous proof would take much space; we will only sketch the method by treating one case. Details can be found in [10]. Examine Figure 1 below.

Before transmission



After collision

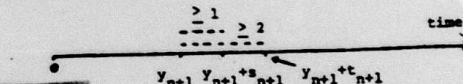


Figure 1: example

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Assume  $s_n \leq F_n < t_n$ , and the transmission results in a collision. Assertion 1 remains true as  $y_{n+1} = y_n$ . To verify assertions 2 and 3, note that due to the Poisson nature of the generation process the numbers of message generations in  $(y_n, y_n + F_n)$  and  $(y_n + F_n, y_n + t_n)$  are independent, and that there are at least two in the first interval implies that there are at least two in the union of the two intervals. Thus the new feedback information that there are at least two generation times in  $(y_n, y_n + F_n)$  makes obsolete the old information that there are at least two in  $(y_n, y_n + t_n)$ , and the only information we have about  $(y_n + F_n, y_n + t_n)$  is the knowledge of its a priori statistics.

Before proceeding with the next section which will show how to define  $F(\cdot, \cdot)$  so as to maximize the rate at which messages are successfully transmitted, we will make two remarks.

First, in its form just given, the algorithm is not causal, in the sense that it sometimes specifies that messages should be transmitted before having been generated. This can be remedied to by defining  $F_n = \min[F(s_n, t_n), n - y_n]$ . However, for the purpose of this paper, we will keep the original form, as we are only interested in the maximum rate at which messages can be successfully transmitted, and not in real time properties, like message waiting times until successful transmission.

Second, this algorithm is the most general algorithm that guarantees that messages are transmitted in the order they were generated (a desirable fairness property), although it is far from being the most general access algorithm.

### III. Analysis and Optimization

The key to the analysis of the algorithm is to realize that the process  $(s_n, t_n)$  is Markovian, as the probabilities of the different outcomes of the  $(n+1)$ th transmission and of the values of  $(s_{n+1}, t_{n+1})$  depend only on  $s_n$  and  $t_n$ . Thus in the case illustrated in the top part of Figure 1, the value of  $(s_{n+1}, t_{n+1})$  will be  $(t_n - F_n, \infty)$  in case of success (i.e., with probability  $\Pr[1 \text{ Poisson arrival in } (y_n, y_n + F_n) | \text{at least 1 Poisson arrival in } (y_n, y_n + s_n) \text{ and at least 2 in } (y_n, y_n + t_n)]$ ). In case of collision the value of  $(s_{n+1}, t_{n+1})$  would be  $(s_n, F_n)$ . The channel cannot remain idle in this case as  $s_n \leq F_n$ .

It is straightforward but tedious to write down the transition probabilities for all cases. We should notice the peculiar role of the  $(\infty, \infty)$  state. Physically it corresponds to all messages generated before  $y_n$  having been successfully transmitted and no information except the a priori statistics being available about generation times greater than  $y_n$ . That state is entered at least every time two consecutive transmissions result in a success, thus it is reachable from all other states.

Moreover if  $F(\cdot, \cdot)$  is such that there is a positive lowerbound on the probability of successful transmission in any state  $(s, t)$  (this is always the case for the  $F(\cdot, \cdot)$ 's considered below), then state  $(\infty, \infty)$  is positive recurrent along with only countably many other states

accessible from it. Thus the computation of state probabilities and expected values, with a given degree of precision, is a straightforward numerical matter.

We will now direct our attention to the problem of selecting  $F(\cdot, \cdot)$  to maximize the long term rate of success (also called throughput), i.e.,

$$\lim_{N \rightarrow \infty} \frac{1}{N} E \left[ \sum_{i=0}^N r(i) \right], \text{ where } r(i) \text{ is equal to one if the } i\text{th transmission is successful, and zero otherwise.}$$

This can be done simply by discretizing the state space and using the successive approximation method (11) of solving undiscounted infinite horizon Markovian decision theory problems.

The details of the work appear in [10]. The following conclusions were reached

- b) the optimal  $F(s, t)$  is never greater than  $s$ , so that all states  $(s, t)$  with  $s \neq t$  or  $t \neq \infty$  are transient.
- c) the optimal  $F(\infty, \infty)$  is  $1.26/\lambda$  so that all states  $(s, t)$  with  $\infty > t \geq s > 1.26/\lambda$  are transient.
- d) the optimal  $F(s, s)$ ,  $s \leq 1.26/\lambda$ , is very close to  $s/2$ . In fact the throughput of the algorithm [9] using  $F(\infty, \infty) = 1.26/\lambda$ ,  $F(s, s) = s/2$ ,  $F(s, \infty) = s$  ( $s < \infty$ ) and  $F(s, \infty) = s$  ( $s < \infty$ ) is .487. This last algorithm is itself a generalization of the tree algorithm [12] which introduced binary splitting.

Note however that remark b) above does not hold for finite horizon (finite  $N$ ) problems, where the optimal  $F(s, \infty)$  at time  $N$  may be larger than  $s$  for  $N \geq 3$  and  $s$  below some threshold, and has a large discontinuity at the threshold. The threshold decreased with  $N$ , becoming smaller than the grid size  $(.01/\lambda)$  for  $N \geq 5$ . No similar behavior was observed for  $F(s, t)$ ,  $t < \infty$ , probably because of the numerical optimization did not consider (transient) states in the region where the phenomenon would occur.

### IV. Unequal Observation Times

In fact many multiaccess communication systems differ from the model introduced in section 1 in that the times necessary to learn the transmission outcomes depend on the outcomes. We denote by  $t_0, t_1$  and  $t_2$  respectively the times necessary to learn that the channel was idle, or that a success or a collision occurred.

For example carrier sense radio systems [2] can detect idles quickly (no carrier present), while they rely on error detecting codes and the transmissions of acknowledgements to distinguish between successes and collisions, thus  $t_0 \ll t_1 = t_2$ . In addition, some cable broadcast systems [3] have a listen-while-transmit feature that allows the quick abortion of transmissions resulting in collisions, thus  $t_0 = t_2 \ll t_1$ .

The general algorithm outlined in section 2 and the remarks about its Markovian nature remain valid, but the reward function  $r(\cdot)$  and the maximization in section III are not appropriate. A better measure of quality is to minimize the expected time to send a message, i.e.,

$$\lim_{N \rightarrow \infty} E \frac{\sum_{i=1}^N \sum_{j=1}^3 t_j I(m(i) = j)}{\sum_{i=1}^N r(i)} = t_1 +$$

$$t_2 \lim_{N \rightarrow \infty} E \frac{\sum_{i=1}^N \frac{t_0}{t_2} I(m(i) = 0) + I(m(i) = 2)}{\sum_{i=1}^N r(i)}$$

where  $m(i)$  is 0,1,2 depending on the outcome of the  $i$ th transmission, and  $I(\cdot)$  denotes the indicator function.

The limit of the expected value in the right hand side can be interpreted as the expected time overhead per message, and depends only on  $t_0/t_2$  for a given  $F(\cdot, \cdot)$ . It will be denoted by  $c$  and should be minimized over  $F(\cdot, \cdot)$  for a given  $t_0/t_2$ .

The optimization of the general algorithm under this formulation is time consuming. It is greatly simplified if we consider only those  $F(\cdot, \cdot)$  such that  $F(s, t) \leq s$ . The only recurrent states are then of the form  $(s, s)$ , or  $(s, \infty)$ , see above. Note that the optimal  $F$  found in section III belonged to the restricted class. We will now show how to proceed with the optimization.

By a renewal argument

$$c = \frac{E \left[ \sum_{i=1}^b \frac{t_0}{t_2} I(m(i) = 0) + I(m(i) = 2) \right]}{E \left[ \sum_{i=1}^b r(i) \right]}$$

where in the right hand side one assumes that  $(s_1, t_1) = (\infty, \infty)$ , and  $b$  is the time of first return at  $(\infty, \infty)$ .

Let us now assume that we guess a value  $\hat{c}$  for the minimum of  $c$  over all restricted  $F(\cdot, \cdot)$ , and consider the function

$$V(s, t) = E \left[ \sum_{i=1}^b \left( \frac{t_0}{t_2} I(m(i) = 0) + I(m(i) = 2) - \hat{c} r(i) \right) \right] \quad |(s_1, t_1) = (s, t)|$$

Because  $s_{n+1}$  is either equal to  $\infty$  or is less than  $s_n$ ,  $V(s, s)$  and  $V(s, \infty)$  can be written as convex combinations of  $V(s', s')$  and  $V(s', \infty)$ ,  $s' < \min(s, F(\infty, \infty))$ . It is straightforward [10] to minimize  $V(s, s)$  and  $V(s, \infty)$  recursively for increasing  $s$ , and to obtain the minimum value of  $V(\infty, \infty)$ .

If the minimum value is 0,  $\hat{c}$  was guessed correctly and is the minimum value of  $c$ . If the minimum value of  $V(\infty, \infty)$  is positive (negative),  $\hat{c}$  was guessed too small (large), and the minimization of  $V(\cdot, \cdot)$  must be repeated with a new  $\hat{c}$ .

The resulting minimum value of  $c$  is shown in Figure 2, as a function of  $t_0/t_2$ . It is almost equal to the expected time overhead per message for the algorithm where the value of  $F(s, s)$  is taken as  $s/2$ ,  $s < \infty$ , the value of  $F(s, \infty)$  as  $s$ ,  $s < \infty$ , and only  $F(\infty, \infty)$  is optimized.

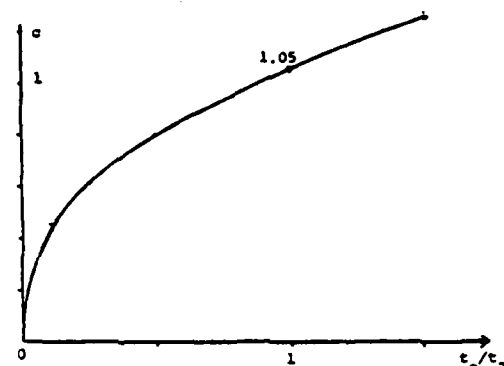


Figure 2:  $c$  for the optimized algorithm

#### V. Noisy Feedback

The previous algorithm assumed that the transmission outcomes were perfectly observed by all sources. This assumption is critical. One verifies easily that if an idle is falsely observed as a collision then the algorithm will deadlock, i.e.,  $y_n \cdot t_n$  will remain constant, while the  $t_n$ 's will decrease to zero.

D. Ryter [13] has recently examined the problem of noisy feedback, where the noise can cause idles or successes to be observed as collisions. He showed that the binary splitting algorithm [9] outlined in section III can be modified to work properly. The essential modification is the introduction of a threshold value. If  $t_n$  is smaller than the threshold, then the algorithm becomes non stationary, in the sense that it alternates between using  $F(s, s) = s$  and  $F(s, s) = s/2$ , thus first seeking confirmation that a collision really occurred, then trying to resolve it. The analysis and optimization are too long to be reported here. The main result is that with the proper choice of parameters, the throughput behaves roughly like  $.467 \cdot p$ , where  $p$  is the probability of false collision indication.

#### VI. Final Comments

The main results of this paper are the description and analysis of an access algorithm for the channel model described in section 1, with infinitely many sources. Its throughput is .488, the largest known to this day. Molle [14] has recently shown that no algorithm can have a throughput higher than .67, and it is widely believed that the best achievable throughput is in the neighborhood of .5. However, throughputs arbitrarily close to 1 are possible, at the expense of high average message delay, when the number of sources is finite.

We have also shown how the algorithm can be modified in the cases of variable transmission times and noisy feedback.

Finally it should be pointed out that although the algorithm presented here uses the message generation times to specify when they should be transmitted, this is not necessary. Another algorithm can be described, with the same throughput and expected time overhead per message, where sources generate random numbers to determine if they should transmit. Of course real time properties, like first-generated-first-transmitted will not be conserved.

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